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Steven I Weisburd Esq			HARPER, V PAUL	
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Please find below and/or attached an Office communication concerning this application or proceeding.

 		Application No.	Applicant(s)				
Office Action Summary		10/046,719	HASEGAWA, SA	HASEGAWA, SATOSHI			
		Examiner	Art Unit				
		V. Paul Harper	2654				
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Status			:				
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3)□	e merits is						
closed in accordance with the practice under Ex parte Quayle, 1935 C.D. 11, 453 O.G. 213.							
Disnositi	on of Claims		:				
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•	Claim(s) <u>1-24</u> is/are pending in the application. 4a) Of the above claim(s) is/are withdraw		,				
	Claim(s) is/are allowed.	wii iioiii consideration	:				
·	Claim(s) <u>1-24</u> is/are rejected.	;					
7) Claim(s) is/are objected to.							
	Claim(s) are subject to restriction and/o	r election requirement	t.				
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Applicati	on Papers		· •				
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10)	The drawing(s) filed on is/are: a)□ acc		- :				
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Priority u	ınder 35 U.S.C. § 119		:				
•	Acknowledgment is made of a claim for foreign ☑ All b)☑ Some * c)⊡ None of:	priority under 35 U.S	.C. § 119(a)-(d) or (f).				
	1. Certified copies of the priority documents have been received.						
	2. Certified copies of the priority document						
	3. Copies of the certified copies of the prior		een received in this Nationa	l Stage			
	application from the International Bureau						
* See the attached detailed Office action for a list of the certified copies not received.							
Attachment	t(s)		· :				
	e of References Cited (PTO-892)		view Summary (PTO-413)				
	e of Draftsperson's Patent Drawing Review (PTO-948) nation Disclosure Statement(s) (PTO-1449 or PTO/SB/08)		r No(s)/Mail Date e of Informal Patent Application (PT	'O-152)			
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DETAILED ACTION

Claim Rejections - 35 USC § 103

1. Claims 1-9, 13, 14, and 16-21 are rejected under 35 U.S.C. 103(a) as being unpatentable over ISO/IEC 11172-3 (as described in the prior art section of the specification, p. 2, Fig. 1 labeled prior art), hereinafter referred to as *Spec_Prior_Art*, in view of Nakajima et al. ("A Fast Audio Classification from MPEG Coded Data" ICASSP '99, vol. 6, May 1999) hereinafter referred to as Nakajima.

Regarding **claim 1**, *Spec_Prior_Art* describes the MPEG1/Audio layer 1 system and includes the following:

- a subband dividing section dividing inputted audio information including a sound signal into a plurality of frequency bands (p. 2, line 15, Fig. 1, item 111);
- a scaling section calculating a scaling factor, which indicates a multiplying power
 to a reference value, of each subband divided by the subband dividing section into
 each of the frequency bands, and aligning each dynamic range (Fig. 1, item 112); and
- a coding processing section compressing and coding an output signal from the scaling section by using a MPEG system to output as coded bit stream data (Fig. 1, items 113-115).

But Spec_Prior_Art does not specifically teach "further including a feature detection processing section extracting features of the audio information on the basis of the scaling factors outputted from the scaling section." However, the examiner contends that this concept was well known in the art, as taught by Nakajima.

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In the same field of endeavor, Nakajima teaches a method for audio classification from MPEG coded data, by processing the sub-band energy levels (§2, the scaling factors necessarily correspond to sub-band energy levels).

Therefore, it would have been obvious to one having ordinary skill in the art at the time the invention was made to modify *Spec_Prior_Art* by specifically providing the features, as taught by Nakajima, because it is well known in the art at the time of invention for the purpose of identifying the content of the audio signal being processed for marketing, monitoring commercials, improved speech recognition (Kenyon et al. U.S. Patent 4,843,562, col. 1), and indexing, browsing, and retrievals from multimedia databases (Nakajima, §1, ¶1).

Regarding **claim 2**, *Spec_Prior_Art* in view of Nakajima teaches everything claimed, as applied above (see claim 1). In addition Nakajima teaches "the feature detection processing section includes a means of determining whether or not the audio information is of a voice signal interval on the basis of the scaling factors" (§1, ¶4, classified into speech; §2.2, "Music/Speech Characteristics" based on the distribution of energy; Fig. 1 and 2, n.b., the amplitude of each histogram corresponds to a subband level).

Regarding claim 3, Spec_Prior_Art in view of Nakajima teaches everything claimed, as applied above (see claim 1). In addition, Nakajima teaches "wherein the feature detection processing section includes a means of determining whether or not

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the audio information is of a soundless signal interval on the basis of the scaling factors" (§2.1, silence, if σ^2 is smaller than the predetermined threshold).

Regarding **claim 4**, this claim has corresponding limitations similar to the limitations in claim 1, and those limitations are rejected for the same reasons. In addition: "a signal level calculating section inputting thereto the scaling factor of each subband outputted from the scaling section, and calculating a signal level corresponding to the scaling factor; wherein the feature detection processing section extracts features of the audio information on the basis of the signal levels calculated by the signal level calculating section" (§2, "Classification Algorithm" Figs. 1 and 2; the amplitude of each histogram corresponds to a scaled sub-band level where this information is used during feature detection).

Regarding **claim 5**, *Spec_Prior_Art* in view of Nakajima teaches everything claimed, as applied above (see claim 4). In addition, Nakajima teaches:

- the signal level calculating section inputs thereto the scaling factors in low-frequency bands outputted from the scaling section within a predetermined period of time to calculate the signal levels (§2, "Classification Algorithm"; §2.1, ¶'s 1 and 2, low frequency; over segment implies a predetermined interval); and
- the feature detection processing section comprises: a calculating means of finding a maximum value and a minimum value of the signal levels calculated by the

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signal level calculating section (§2, certain level of variation; requires the determination of min/max--range), and

- calculating a difference between the maximum value and the minimum value (§2, variation);
- a determining means of, when the difference value calculated by the calculating means is greater than or equal to a predetermined threshold value, determining that the audio information is of a voice signal interval, on the other hand, when the difference value is less than the threshold value, determining that the audio information is of a signal interval except for voice (§2, silence, if σ^2 is smaller than the predetermined threshold, otherwise the audio information is evaluated for speech, etc).

Regarding **claim 6**, *Spec_Prior_Art* in view of Nakajima teaches everything claimed, as applied above (see claim 4). In addition, Nakajima teaches:

- the signal level calculating section inputs thereto all of the scaling factors
 outputted from the scaling section within a predetermined period of time to calculate
 the signal levels (§1, from MPEG coded data; see Figs, 1 and 2); and
- the feature detection processing section includes a determining means of, when the signal levels calculated by the signal level calculating section are greater than or equal to a predetermined threshold value (§2, silence if σ^2 is smaller than the predetermined threshold, otherwise the audio information is evaluated for speech, etc),
- determining that the audio information is of a sound signal interval (§2.2
 "Music/Speech Characteristics", not silence),

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• on the other hand, when the signal levels are less than the threshold value, determining that the audio information is of a soundless signal interval (§2, silence, if σ^2 is smaller than the predetermined threshold).

Regarding **claim 7**, this claim has limitations similar to claim 1 and is rejected for the same reasons.

Regarding **claim 8**, this claim has limitations similar to claim 2 and is rejected for the same reasons.

Regarding **claim 9**, this claim has limitations similar to claim 3 and is rejected for the same reasons.

Regarding **claim 13**, this claim has limitations similar to claim 1 and is rejected for the same reasons.

Regarding **claim 14**, this claim has limitations similar to claim 2 and is rejected for the same reasons.

Regarding **claim 16**, this claim has limitations similar to claim 4 and is rejected for the same reasons.

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Regarding **claim** 17, this claim has limitations similar to claim 5 and is rejected for the same reasons.

Regarding **claim 18**, this claim has limitations similar to claim 6 and is rejected for the same reasons.

Regarding **claim 19**, this claim has limitations similar to claim 7 and is rejected for the same reasons.

Regarding **claim 20**, this claim has limitations similar to claim 8 and is rejected for the same reasons.

Regarding **claim 21**, this claim has limitations similar to claim 9 and is rejected for the same reasons.

2. Claims 10-12, 15, and 22-24 are rejected under 35 U.S.C. 103(a) as being unpatentable over *Spec_Prior_Art* in view of Nakajima and well known prior art (MPEP 2144.03).

Regarding **claim 10**, *Spec_Prior_Art* describes the encoding portion of ISO/IEC 11172-3, but does not specifically describe "a stream dividing section, after inputting thereto bit stream data coded by a MPEG system, dividing the coded bit stream data

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composed of each subband divided into each frequency band into bit assigning information, scaling factor value indicating a multiplying power to a reference value, and coded data in units of each subband; and a decoding processing section executing a decoding process to the coded data divided by the stream dividing section in units of each subband to output as audio information." However, the examiner takes official notice of the fact that the use of a decoder for the purpose of decoding data encoded according to ISO/IEC 11172-3 was well known in the art.

Therefore, it would have been obvious to one having ordinary skill in the art at the time the invention was made to modify *Spec_Prior_Art* such that a decoder is implemented, because a decoder is part of the ISO/IEC 11172-3 specification and required for the complete processing of the signal.

In addition, Spec_Prior_Art does not specifically teach:

- a feature detection processing section extracting features of the audio
 information on the basis of the scaling factor values outputted from the stream dividing section; and
- a signal level calculating section inputting thereto the scaling factor of each subband outputted from the stream dividing section to calculate a signal level;
- wherein the feature detection processing section extracts features of the audio information on the basis of the signal levels calculated by the signal level calculating section.

However, the examiner contends that these concepts were well known in the art, as taught by Nakajima.

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In the same field of endeavor, Nakajima teaches a method for audio classification from MPEG coded data, where Nakajima processes the sub-band energy levels (§2, where the scaling factors necessarily correspond to sub-band energy levels since Nakajima is processing sub-band energy levels) and performs classification (§2, silence, speech, etc.).

Therefore, it would have been obvious to one having ordinary skill in the art at the time the invention was made to modify $Spec_Prior_Art$ by specifically providing the features, as taught by Nakajima, because it is well known in the art at the time of invention for the purpose of identifying the content of the audio signal being processed for marketing, monitoring commercials, improved speech recognition (Kenyon et al. U.S. Patent 4,843,562, col. 1), and indexing, browsing, and retrievals from multimedia databases (Nakajima, §1, ¶1).

Regarding **claim 11**, *Spec_Prior_Art* in view of Nakajima and well known prior art teaches everything claimed, as applied above (see claim 10). In addition, Nakajima further teaches:

• the signal level calculating section inputs thereto the scaling factors in low-frequency bands outputted from the stream dividing section within a predetermined period of time to calculate the signal levels (§2, "Classification Algorithm" most of the sub-band energy is confined to the lower sub-bands and variations are compared); and

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• the feature detection processing section comprises: a calculating means of finding a maximum value and a minimum value of the signal levels calculated by the signal level calculating section (§2, variation is determined), and

- calculating a difference between the maximum value and the minimum value (§2, variation is determined with a necessary calculation of min/max difference); and
- a determining means of, when the difference value calculated by the calculating means is greater than or equal to a predetermined threshold value, determining that the audio information is of a voice signal interval, on the other hand, when the difference value is less than the threshold value, determining that the audio information is of a signal (§2.1, silence, if σ^2 is smaller than the predetermined threshold).

Regarding **claim 12**, *Spec_Prior_Art* in view of Nakajima and well known prior art teaches everything claimed, as applied above (see claim 10). In addition, Nakajima further teaches:

- the signal level calculating section inputs thereto all of the scaling factors
 outputted from the stream dividing section within a predetermined period of time to
 calculate the signal levels (§2, time and frequency analysis, frames in one second);
- the feature detection processing section includes a determining means of, when the signal levels calculated by the signal level calculating section are greater than or equal to a predetermined threshold value (§2.1, silence if σ^2 is smaller than the predetermined threshold);

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• determining that the audio information is of a sound signal interval (§2, §2.1, "Silence Segment Detection" if not silence necessarily "sound"),

determining that the audio information is of a sound signal interval, on the other hand, when the signal levels are less than the threshold value, determining that the audio information is of a soundless signal interval (§2.1, silence, if σ^2 is smaller than the predetermined threshold).

Regarding **claim 15**, this claim has limitations similar to claim 3 and is rejected for the same reasons.

Regarding **claim 22**, this claim has limitations similar to claim 10 and is rejected for the same reasons.

Regarding **claim 23**, this claim has limitations similar to claim 11 and is rejected for the same reasons.

Regarding **claim 24**, this claim has limitations similar to claim 12 and is rejected for the same reasons.

Citation of Pertinent Art

3. The following prior art made of record but not relied upon is considered pertinent to the applicant's disclosure:

- Zick et al. (U.S. Patent 6,370,504) disclose an invention that performs speech recognition on MPG/Audio encoded files.
- Patel et al. ("Audio Characterization for Video Indexing" SPIE, 1996, pp. 373-384) teaches audio feature extraction from an MPEG encoded audio stream.

Conclusion

Any inquiry concerning this communication or earlier communications from the examiner should be directed to V. Paul Harper whose telephone number is (571) 272-7605. The examiner can normally be reached on M-F.

If attempts to reach the examiner by telephone are unsuccessful, the examiner's supervisor, Richemond Dorvil can be reached on (571) 272-7602. The fax phone number for the organization where this application or proceeding is assigned is 571-273-8300.

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Information regarding the status of an application may be obtained from the Patent Application Information Retrieval (PAIR) system. Status information for published applications may be obtained from either Private PAIR or Public PAIR. Status information for unpublished applications is available through Private PAIR only. For more information about the PAIR system, see http://pair-direct.uspto.gov. Should you have questions on access to the Private PAIR system, contact the Electronic Business Center (EBC) at 866-217-9197 (toll-free).

1. Paul Horper

12/29/2005

V. Paul Harper Patent Examiner Art Unit 2654